

A RECEIVER DEVICE FOR A MOBILE RADIOCOMMUNICATION  
UNIT EMPLOYING A SPEED ESTIMATOR

CROSS-REFERENCE TO RELATED APPLICATIONS

5 This application is based on French Patent Application No. 00 11 118 filed August 31, 2000, the disclosure of which is hereby incorporated by reference thereto in its entirety, and the priority of which is hereby claimed under 35 U.S.C. §119.

BACKGROUND OF THE INVENTION

Field of the invention

10 The present invention relates to a receiver device for a mobile radiocommunication unit employing a speed estimator.

The invention relates more particularly to the field of telecommunications and especially to the field of radiocommunication terminals.

Description of the prior art

15 In radiocommunication terminals, the signals received by the receiver of a mobile receiver unit are degraded because of propagation channel variations. The propagation channel variations depend mainly on the speed of the mobile receiver unit. The channel variations lead to a channel estimation error. The unwelcome consequence of this is that the bit error rate is significantly degraded when the received signal is decoded. Also, a propagation channel estimator can be provided in  
20 the structure of the receiver of the radiocommunication terminal in order to take account of amplitude variations of the signal received by the antenna of the receiver due to the speed of the mobile receiver unit.

25 However, the propagation channel estimator is insufficient for determining the impulse response of the channel with good accuracy. When the speed of the mobile receiver unit increases, the propagation channel varies too quickly for the propagation channel estimator to be able to estimate the frequency and phase variations with sufficient accuracy.

30 An alternative set out in the patent document GB 2 276 064 consists of using Wiener filtering in the receiver. A Wiener filter is a digital filter with a finite impulse response. The amplitude of the output signal of a Wiener filter is closely related to that of the input signal. In other words, a Wiener filter is a filter in which the output signal at a given time depends only on the input signal at that time.

35 To alleviate the problem of propagation channel variation and the resulting degraded receiver signals, the patent previously cited discloses the use of a plurality of Wiener filters, each set for a range of contiguous speeds of the mobile receiver



as a function of the speed of the mobile radiocommunication unit, which receiver device further includes a speed estimator for estimating the speed of the mobile radiocommunication unit, whose input is connected to the output of the channel estimator and whose output is connected to a second input of the filter unit, thereby  
 5 supplying to it the estimated speed of the mobile radiocommunication unit in order to select the appropriate Wiener filter corresponding to the estimated speed.

The invention also provides a method of estimating the speed of a mobile radiocommunication unit in a receiver device communicating with a base station via a propagation channel, which method consists in estimating the speed by measuring  
 10 the phase difference between two channel coefficients obtained from a channel estimator in accordance with the following equation:

$$V_{n,p} = c \cdot (\phi_{n+p} - \phi_n) / 2\pi \cdot f_c \cdot T_s$$

in which:

$V_{n,p}$  is the speed at time  $n$ , calculated with a difference  $p$  between the two  
 15 phases of the two channel coefficients taken into consideration,

$c$  is the speed of light;

$f_c$  is the carrier frequency,

$T_s$  is the sampling period of the channel coefficients,

$\phi_n$  is the phase of the channel coefficient at time  $n$ , and

20  $\phi_{n+p}$  is the phase of the channel coefficient at time  $n+p$ .

Other features and advantages of the invention will become more clearly apparent on reading the following description of one particular embodiment, which is given with reference to the drawings.

#### BRIEF DESCRIPTION OF THE DRAWINGS

25 Figure 1 is a diagrammatic representation of a rake receiver in a mobile radiocommunication system in one embodiment of the present invention.

Figure 2 is a diagram representing for each path the amplitude of the power of the received signal as a function of the time-delay.

Figure 3 shows the mechanism for assigning Wiener filters in a preferred  
 30 embodiment of the invention.

Figure 4 shows performance in terms of bit error rate as a function of signal/noise ratio.

Figure 5 shows the variations in the difference between two phases to be measured as a function of the speed of the mobile receiver unit.

35 Figure 6 shows variations as a function of time of the time constant of a low-

pass filter used in the method according to the invention.

#### DETAILED DESCRIPTION OF THE DRAWINGS

Figure 1 therefore shows diagrammatically an embodiment of the present invention in which a mobile telecommunication system uses a rake receiver.

5 A base station 1 transmits signals in all directions to all radiocommunication terminals inside its coverage area. The radio waves are transmitted via a propagation channel 2. The propagation channel 2 corresponds to the path followed by the radio waves between their point of transmission and their point of reception. The signals transmitted are affected by Gaussian additive white noise 3. The amplitude and time-  
10 delay values of the impulse response of the propagation channel 2 are a function in particular of the environment, i.e. of the region of the globe concerned. The processing of the external noise 3 is modeled by an adder 4 in which the signal from the propagation channel 2 is added to the external noise 3. The signal modeled in this way reaches a receiver device 5.

15 The signal modeled in this way includes the wanted signal and takes account of the external noise. It is applied to the single input of a pathfinder circuit 6, to a first input of a channel estimator 7, and to a first input of a combiner circuit 10. The pathfinder circuit 6 has an output connected to a second input of the speed estimator 7 and to a second input of the combiner circuit 10.

20 The channel estimator 7 has an output connected, on the one hand, to a first input of a filter unit 9 which, in a preferred embodiment of the invention, is made up of a plurality of Wiener filters, i.e. constitutes a Wiener filter bank, and, on the other hand, to a single input of a mobile receiver unit speed estimator 8. The speed estimator 8 has an output connected to a second input of the filter unit 9 consisting of  
25 a plurality of Wiener filters.

The filter unit 9 has an output connected to a third input of the combiner circuit 10.

30 The pathfinder circuit 6 cooperates with the channel estimator 7 to determine the profile of the transmission channel in terms of its time-delay, phase and amplitude.

In fact, the device according to the invention is part of a radiocommunication system employing multipath propagation. The radio signal therefore propagates along one or more paths, one of which is the shortest path connecting the point of transmission, the base station 1, to the point of reception, the  
35 receiver 5, and the others of which are due to obstacles from which the waves

ricochet before reaching the receiver 5 with phases different from that of the wave that took the shortest path. The reflected waves travel distances different from that traveled by the direct wave and their phases therefore lag relative to the phase of the direct wave.

5           What is more, the waves arriving with a time-delay have taken a longer path and are consequently more attenuated, which means that their amplitudes are different.

          The signal therefore reaches the mobile receiver unit with phase and amplitude distortion.

10           The function of the pathfinder circuit 6 is then to estimate the time-delays in the transmission of the signals due to the multipath phenomenon explained hereinabove. To do this, the circuit 6 deduces the time-delays from a power estimate for each path. The pathfinder circuit 6 receives at its input the multipath signal and delivers at its output, after processing in a manner that is known in the art, using  
15           various algorithms, the power profile of the signal over a certain time, as shown in figure 2. The circuit 6 uses in particular means for correlating the pilot sequence of the mobile receiver unit with the received signal.

          Figure 2 is a diagram representing for each path the amplitude of the powers of the received signals as a function of the time-delays. The time-delays are  
20           plotted on the abscissa axis and for each time-delay value  $\tau_1, \tau_2, \tau_3, \tau_4, \tau_5, \dots, \tau_i$  there is a corresponding power amplitude plotted on the ordinate axis. The signals represented convey the same information, and simply arrive at the receiver with time, phase and amplitude differences. Usually, the greater the time-delay on a given path, the greater the attenuation of the power amplitude of the signal received at the  
25           receiver. For example, the power of the received signal has a lower amplitude for the path that has a cumulative time-delay  $\tau_i$  relative to the first path, which means that the wave  $i$  has taken a long path and/or been subjected to attenuation due to the environment before reaching the mobile receiver unit. These paths are not taken into account hereinafter. In fact, by means of other algorithms, a decision is taken to fix a  
30           particular threshold and to retain only paths that have a power level greater than the noise, i.e. those which must be used to maintain communication between the base station 1 and the receiver 5.

          Once the various time-delays have been determined, by means of the processing carried out by the pathfinder circuit 6, the channel estimator 7 comes into  
35           play and supplies a first estimate of the impulse response of the propagation channel.

In other words, the function of the channel estimator 7 is to determine the amplitude and the phase of each path. To meet this objective, it is necessary for the input of the channel estimator 7 to receive the multipath signal and the time-delays calculated by the pathfinder circuit 6. The values of the time-delays for the various paths  $\tau_1, \tau_2, \tau_3, \tau_4, \tau_5, \dots, \tau_i$ , as discussed with reference to figure 2, must therefore be supplied to the channel estimator 7 by the pathfinder circuit 6. In fact, the channel estimator 7 must know the value of the time-delay  $\tau$  of each path in order to be able to determine the amplitude and the phase of the signal for each path.

Based on the above data, the channel estimator 7 knows that there is a path at  $\tau_1$ , at  $\tau_2$ , . . . at  $\tau_i$ . It then calculates the amplitude and the phase of the multipath signal at times  $\tau_1, \tau_2, \dots, \tau_i$ . The amplitude and the phase for each path are then represented by a coefficient.

The amplitude and phase coefficients are then supplied to the speed estimator 8. The speed estimator 8 uses the path coefficients calculated by the channel estimator 7 to estimate the speed of the mobile receiver unit. In a different embodiment, the speed estimator could use the path having the highest power to estimate the speed of the mobile unit, rather than using all the paths.

The amplitude and phase coefficients have a phase that varies as a function of the Doppler effect due to the speed of the mobile receiver unit. The speed estimator 8 therefore measures the phase variation, which is closely related to the speed of the mobile receiver unit. The operation of the speed estimator is described in detail next with reference to figures 5 and 6 in particular.

The speed estimator 8 then supplies the estimate of the speed of the mobile receiver unit to the unit 9 which, in this embodiment, comprises a plurality of Wiener filters. The most appropriate Wiener filter coefficients are deduced from the speed estimate. In fact, there is one filter that corresponds to each speed. The objective of the Wiener filtering is to filter the channel coefficients.

There are various ways to assign a filter as a function of the speed. In theory, a suitable Wiener filter would be required for each speed. However, this kind of solution would entail long calculations to discover the filter exactly matched to the speed and would therefore be costly in terms of processing time.

A bank of Wiener filters is then used, in which each filter is matched to a different range of contiguous speeds. A particular filter is used when the speed of the mobile receiver unit is inside a predetermined speed range.

Figure 3 shows the mechanism for assigning the Wiener filters in a preferred



The error rate represents the percentage of errors in the digital signal received by the mobile receiver unit.

In figure 4, a first curve S1 corresponds to the performance in terms of the BER as a function of the ratio  $E_b/N_0$  when a Wiener filter whose parameters have been set by the speed estimator, i.e. in accordance with the present invention, is used in the mobile receiver unit. A second curve S2 corresponds to performance in terms of the BER as a function of the ratio  $E_b/N_0$  if an ideal filter matched to the exact speed of 37.5 kilometers per hour is used in the mobile receiver unit. The curves S1 and S2 are the same. Finally, a curve S3 corresponds to performance in terms of the BER as a function of the ratio  $E_b/N_0$  if no Wiener filter is used.

A BER of  $10^{-3}$  is considered by way of example. A BER of  $10^{-3}$  means that the required quality of service corresponds to one wrong data bit every thousand bits.

For a BER of  $10^{-3}$ , the signal/noise ratio  $E_b/N_0$  for the curve S1, representative of the situation in which a Wiener filter whose parameters have been set by the speed estimator is used, is 7.2 decibels. For the curve S2, representative of the ideal filter, the ratio  $E_b/N_0$  is also 7.2 decibels for a BER of  $10^{-3}$ . Thus, by using a Wiener filter whose parameters are set by the speed estimator, the same performance is obtained as with the ideal filter.

In contrast, for the curve S3, representative of the situation in which no Wiener filter is used, the ratio  $E_b/N_0$  is 7.7 decibels, i.e. 0.5 decibels worse than for the curve S1. Thus, in this case, to obtain the same quality of service, it is necessary to provide a higher base station and mobile transmit power.

Using in the mobile receiver unit a Wiener filter whose parameters are set by the speed estimator, i.e. in accordance with the present invention, obtains a power saving of 0.5 decibels at 37.5 kilometers per hour and therefore enables the base station to transmit at a lower power. This phenomenon has a particular importance in the context of the universal mobile telecommunication system (UMTS) standard, in accordance with which the number of users for a base station is intimately related to the transmit power. Accordingly, the lower the transmit power, the greater the number of users for the same base station.

The speed estimator is described in more detail next with reference to figures 5 and 6 in particular. The speed is estimated by means of a simple process that is suitable for any type of propagation channel.

To improve the quality of service it is very important to know the speed of the mobile receiver unit. In fact, the speed of the mobile receiver unit causes channel



variations and this has a direct impact on the channel estimate and consequently on the bit error rate BER. To improve reception quality, a channel estimate is employed, followed by Wiener filtering of the impulse response of the propagation channel. However, if high processing performance is to be obtained, the Wiener filter to be used must be matched to the speed of the mobile receiver unit. This is why, in accordance with the invention, a speed estimator is used in the receiver device to set the parameters of the Wiener filter to be used.

The method according to the invention is based on the principle of the Doppler frequency which, as is well known, is related to the speed of the mobile receiver unit. The speed of the mobile receiver unit is related to the propagation channel variations, which variations cause distortion of the signal, in particular phase variation.

Accordingly, the method of estimating the speed consists in measuring the Doppler frequency by calculating the phase difference between two channel coefficients. The method in accordance with the invention uses the channel impulse response from the channel estimator to measure the phase difference between the impulse responses of two channels. Equation 1 below shows the relation between the channel estimate phase difference and the speed of the mobile receiver unit:

$$V_{n,p} = c. (\phi_{n+p} - \phi_n) / 2\pi.f_c.T_s \quad \text{equation 1}$$

in which:

$V_{n,p}$  is the instantaneous speed at time  $n$ , calculated with a phase difference  $p$  between the phases of the two channel estimates taken into consideration;

$c$  is the speed of light;

$f_c$  is the carrier frequency, of the order of 2 GHz in a UMTS system;

$T_s$  is the sampling period of the channel coefficients and in this example represents 666 microseconds;

$\phi_n$  is the phase of the channel coefficient at time  $n$ ; and

$\phi_{n+p}$  is the phase of the channel coefficient at the time  $n+p$ .

To estimate the speed in this way, it is therefore necessary first to store channel coefficients from the channel estimator.

A first step of the speed estimation method in accordance with the invention consists in adaptive measurement of the speed as a function of the power profile of the multipath signal, as shown in figure 2. Indeed, when the signal/noise ratio  $E_b/N_0$  is too low, the signal cannot be distinguished from noise. The speed measurement is then not representative and may be totally erroneous.

To estimate a representative speed of the mobile receiver unit, a speed is measured for each path  $i$ , as shown in figure 2. This measurement of the speed on each path is performed in accordance with equation 1. All the paths can be taken into account, or just a few paths.

- 5 A final estimate of the speed is then obtained by weighting the estimated speed on each path as a function of the power. The various speeds are therefore combined as a function of the power profile of the multipath signal, in accordance with the following equation:

$$\hat{V}_{n,p} = \frac{\sum_{i=1}^N \hat{V}_{n,p,i} \cdot \alpha_i}{\sum_{i=1}^N \alpha_i}$$

in which:

- 10 -  $\hat{V}_{n,p}$  is an estimate of the instantaneous speed obtained by means of the speeds  $\hat{V}_{n,p,i}$  extracted from the measurements performed on the various paths  $i$ , and

-  $\alpha_i$  are coefficients between 0 and 1, calculated as a function of the amplitude of the power of each path  $i$ .

- 15 To calculate the coefficients  $\alpha_i$ , the average power can be measured on each path with a first order filter. Accordingly, each coefficient  $\alpha_i$  is calculated as a function of the average power  $P_{i,avg}$  and the instantaneous power  $P_{i,inst}$  of the path  $i$ . If the instantaneous power is below a particular threshold relative to the average power, the corresponding estimated speed is not taken into account.

- 20 In this first step, the speed is therefore estimated taking the power profile of the multipath signal into account.

Estimating the speed on each path entails several operations.

- 25 Accordingly, a second step consists in estimating the phase variation and adapting  $p$  as a function of the speed of the mobile. The value  $p$  corresponds to the difference expressed as the number of samples between the two phases to be measured to calculate the phase difference. One is taken at time  $n$  and the other at time  $n+p$ . Varying  $p$  as a function of the speed of the mobile enables the phase variation to be calculated under all circumstances, regardless of the Doppler frequency variation.

- 30 Figure 5 shows the variation of  $p$  as a function of the speed  $V$ . The value of

p is between  $p_{\min}$  and  $p_{\max}$ , depending on the value of the speed. If the speed  $V$  is low,  $p$  is large and is equal to  $p_{\max}$ ; thereafter, as the speed increases,  $p$  decreases to the value  $p_{\min}$ .

5 The value of  $p$  can be calculated using a linear function of the type  $p = A.V_{n,p} + B$ , in which  $A$  and  $B$  are constants and  $p$  is an integer. This relation between  $p$  and the speed must be seen as an example and is in no way limiting on the invention. Any other equation establishing a variation of  $p$  as a function of the speed can be envisaged.

10 Accordingly, measuring the speed using two channel coefficients spaced by a number of samples equal to  $p$ , where  $p$  is matched to the speed, reduces the average estimation error due to the Gaussian additive white noise operative on the channel coefficients. The following equation shows the reduction of the average error:  
 $\hat{V}_{n,p} = V_{n,p} + K.(\varepsilon + n) / p$

in which:

- 15 -  $\varepsilon$  is the average estimation error dependent on the speed;  
-  $n$  is Gaussian noise;  
-  $K = c / (2\pi.f_c.T_s)$ ; and  
-  $\hat{V}_{n,p}$  is the estimate of the speed at time  $n$  and  $V_{n,p}$  the real speed. The speed estimate is therefore equal to the real speed plus a certain error.

20 Here  $p$  is a divisor of the average error. Dividing by  $p$  reduces the average error.

A third step consists in calculating the instantaneous speed in accordance with equation 1 above.

25 A fourth step consists in averaging the estimates of the instantaneous speed using a filter to limit the noise  $n$ . In a preferred embodiment of the invention, this filtering can be applied by means of a low-pass filter with time constant  $\delta$ .

Finally, a fifth step improves the convergence of the algorithm when it is launched.

30 Figure 6 shows this fifth step. Figure 6 shows the variations in the time constant  $\delta$  of the low-pass filter used in the preceding step as a function of time, to be more precise as a function of time slices. In fact, as explained above, the channel coefficients from the channel estimator are discrete and therefore sampled with an increment  $T_s$ .

35 The time constant  $\delta$  therefore varies between  $\tau_{\min}$  and  $\tau_{\max}$ . Initially, the time constant  $\delta_{\min}$  is low, which corresponds to fast convergence, enabling the value

of the average speed step to be achieved fairly quickly. Thereafter, the value of  $\delta$  increases as a function of the time slices up to the value  $\delta_{\max}$ , thereby reducing fluctuations due to noise.

- Dynamic management of the time constant of the instantaneous speed filter
- 5 therefore improves the convergence of the algorithm in a number of time slices.